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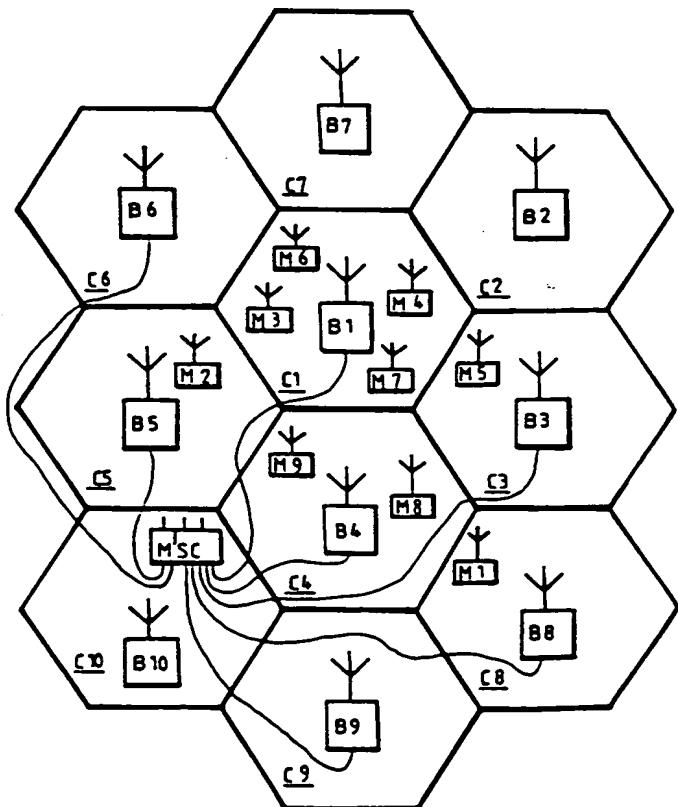
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(54) Title: METHOD OF MAINTAINING AN ESTABLISHED CONNECTION IN A MOBILE RADIO SYSTEM COMPRISING BOTH ANALOG AND DIGITAL RADIO CHANNELS

(57) Abstract

The invention relates to a method in a mobile radio system having both digital and analog radio channels. An object of the invention is that a connection established on a digital radio channel often shall be possible to maintain even when the time dispersion on available digital radio channels exceeds the designed maximum time dispersion of equalizers in receiving stations. According to the invention equalizers in the receivers of the system may be designed for a smaller maximum time dispersion than the total time dispersion the whole system shall be capable to handle. Instead handover is made to an other channel when the time dispersion of a used digital channel tends to become too big. When another digital radio channel with low time dispersion is available handover is preferably done to this channel. Alternatively or in the absence of such a digital radio channel handover is performed to an analog radio channel.



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METHOD OF MAINTAINING AN ESTABLISHED CONNECTION IN A MOBILE RADIO SYSTEM COMPRISING BOTH ANALOG AND DIGITAL RADIO CHANNELS

FIELD OF INVENTION

The present invention relates to the technical field of mobile radio systems. More specifically the invention relates to a method in a mobile radio system having both analog and digital radio channels. The purpose of the invention is that a connection which has begun on a digital radio channel often shall be possible to sustain even when the dispersions on available digital radio channels exceed the maximum time dispersion for which the equalizers in the mobile stations have been designed. Due to its character the method might be referred to the technical field of handover in mobile radio systems.

BACKGROUND OF THE INVENTION

The mobile radio systems first in common use had analog transmission of information between base stations and mobile stations, i.e. analog information was transmitted between base and mobile stations by using analogously-modulated radio signals on analog radio channels. By frequency multiplexing technique analog mobile radio systems may have plural radio channels. The Nordic mobile telephone system, NMT, is an example of such an analog mobile radio system having many radio channels.

Recently there has been suggested mobile radio systems with digital transmission of information between base stations and mobile stations, i.e. digital or digitized information is transmitted between base and mobile stations using digitally modulated radio signals on digital radio channels. Of cause more radio channels may be created by frequency multiplex in digital mobile radio systems also. In order to get even more channels in digital mobile radio systems it has been proposed that digital radio channels shall share a radio frequency in time multiplex. The pan european mobile radio system, GSM, may be mentioned as an example of such a digital mobile radio system.

The digital technique means substantial advantages in mobile radio systems. Accordingly there is an interest in introducing the digital technique. On the other hand there are in certain areas existing analog mobile radio systems using the frequency band available for a digital system. When huge investments have been made in existing analog systems it is not always reasonable to immediately replace an existing analog system with a digital system. It has therefore been suggested to successively replace analog channels in analog systems having many analog channels in frequency multiplex by a number of digital channels in time multiplex. For USA it has been suggested that each analog channel shall be replaced by three digital channels in time multiplex. During a long transitory period of time such a mobile radio system would then comprise both analog radio channels for analog transmission of information by using analog-modulated radio signals and digital radio channels for digital transmission of information by using digitally modulated radio signals. Mobile stations which during the transitory time wish to be able to use the full capacity of such a system must then be able to use both analog radio channels and digital radio channels.

In digital mobile radio systems problems occur with time dispersion on the digital radio channels due to reflexions and multipath propagation unless particular measures are taken. The time dispersion changes from one place to the other due to the radio propagation properties. In order to obtain a reliable transmission of information in spite of time dispersion it is well known to have equalizers at the receiving side in a mobile radio system. Depending upon design such an equalizer can handle shorter or longer time dispersion. One way to describe the ability of an equalizer to cope with time dispersion is to state the maximum time dispersion in microseconds the equalizer is designed for. Another way is to state how many symbol time spans the equalizer can cope with. The complexity, cost and power consumption of an equalizer increases progressively with the maximum time dispersion in microseconds and the number of symbol duration times. Accordingly there is an interest in designing a mobile radio system such that the requirements on the equalizers do not become to

great and in particular not become greater than necessary. For the above mentioned pan european system GSM it has been decided that the equalizers shall be designed to cope with a maximum time dispersion of 16 microseconds, which corresponds to four symbol time intervals. In the above mentioned mobile radio system for USA having both digital and analog radio channels the system should probably need to cope with a maximum time dispersion of about 60 microseconds on the digital radio channels, which is more than the duration of one symbol but less than the duration of two symbols.

The analog transmission of information on the analog radio channels of existing analog mobile radio systems, e.g. the above mentioned NMT or of the type existing in the USA, is normally substantially less sensitive than digital systems to reflexions and multipath propagation causing dispersion on the digital radio channels.

BRIEF DESCRIPTION OF THE INVENTION

The complexity, cost and current consumption of equalizers increase progressively with the maximum time dispersion that an equalizer shall be able to handle or the number of symbol duration times the equalizer shall handle. This causes problems when designing mobile radio systems where a long time dispersion must be considered. The problems become particularly great when for various reasons it is not possible to choose certain parameters more or less freely, e.g. when desiring to successively change from analog technique to digital technique in a mobile radio system by successively replacing analog radio channels by time multiplexed digital radio channels. It can become particularly unfavourable if the time dispersion the system ought to successfully handle is substantially greater than a certain integer number of symbol time intervals but at the same time substantially less than the closest succeeding higher integer number of symbol time intervals. If then desiring to cope with the entire time dispersion by an Viterbi equalizer it is necessary to design the equalizer for the biggest one of the two integer numbers of symbol duration. The invention

aims at solving this problem in a mobile radio system having both analog and digital radio channels.

Somewhat simplified one could say that the invention is based upon the idea to use the lower sensity of the analog radio channels for reflexions and multipath propagation causing dispersion on the digital radio channels. The invention is also based on the understanding that the time dispersion on a particular radio channel in most cases is substantially less than the maximum time dispersion the system is to be designed for. Somewhat simplified it could the be said that according to the invention the equalizers are designed for a lower maximum time dispersion corresponding to the lower of the integer numbers of symbol time intervals and instead at least a temporary change over from a digital radio channel to an analog radio channel is done when the time dispersion on the used digital radio channel becomes greater than the maximum time dispersion the equalizer is designed for. Accordingly handover is done to an available analog radio channel. If there should be another available digital radio channel being sufficiently good the first choice is to attempt to use this by handover to such a digital radio channel. In such cases handover to an available analog radio channel is done only when there is no sufficiently good available digital radio channel.

It is possible but not always self evident to return to a digital radio channel by handover as soon as there is an available digital radio channel being sufficiently good.

In order to get to know in time when the dispersion on the used digital radio channel tends to become too severe the radio propagation properties of the used digital radio channel ought to be estimated more or less continuously. The estimation is preferably done by estimation of signal level and symbol error frequency, which can be performed in ways known per see. One method has been presented by Italtel within the framework on the pan european GSM system and is disclosed in a GSM paper titled GSM/WP2 doc 17/88. Based on the method suggested by Italtel Ericsson has suggested a method of measuring and transferring bit error rate information

for the future U.S. digital cellular standard in a paper forwarded to the TIA Technical Subcommittee, TR-45 Digital Cellular Standards, meeting on August 29, 1989 in San Diego, California.

5 The estimation of signal level and symbol error frequency is preferably performed at the mobile station which reports the estimation to its base station. This is in accordance both with the GSM standard and the TIA on digital cellular mobile radio systems.

10 Instead of estimating time dispersion indirectly by estimation of bit error rate and signal level the time dispersion may be estimated directly using receiver equalizer and synchronizing words transmitted on digital radio channels. Such direct estimation may use methods based on the contribution to the 38th IEEE vehicular technology conference held in Philadelphia, Pennsylvania, USA on 15 15-17 June, 1988, titled "Bit synchronization and timing sensitivity in adaptive viterbi equalizers for narrowband-TDMA digital mobile radio systems", by A. Baier, G. Heinrich, U. Wellens.

20 When possible it is preferred to perform a more or less continuous estimation also for at least one other digital radio channel than the channel presently used by the mobile. At least the radio propagation properties of at least one other possible digital radio channel shall be performed when the estimation of the radio propagation properties of the used digital radio channel indicates that the used digital radio channel is or tends to be too poor.

25 In order to increase the likelihood of successful handover to an analog radio channel the properties of the intended analog radio channel also ought to be estimated in an appropriate way, e.g. the expected signal level and/or the signal to noise ratio of the channel may be estimated. When this is difficult to accomplish at 30 the mobiles it may be done at the base stations.

When introducing digital radio channels into a known mobile radio system with analog radio channels where decisions on handoff are done in the fixed part of the system and estimation of dispersion

and other radio signal properties may also be done in the fixed part of the system the normal known handoff signalling procedure of the system for analog radio channels may be used. In a mobile radio system with a mobile assisted handoff procedure where at least some measurements or estimations are done at the mobile the procedure for handoff from a digital radio channel to an analog radio channel due to dispersion must include some signalling of results of measurements or estimations from mobile to base station.

10 THE DRAWINGS

Figure 1 illustrates part of a cellular mobile radio system with cells, a mobile switching centre, base stations and mobile stations.

15 Figure 2 illustrates a plurality of radio channels within a frequency band and use of some radio channels in a cellular mobile radio system according to figure 1.

20 Figure 3 illustrates use of radio channels according to figure 2 for control channels, analog communication channels and time division multiplex digital communication channels in a cellular mobile radio system according to figure 1.

Figure 4 illustrates bursts separated by guard spaces on a radio channel used for digital communication channels in time division multiplex according to figure 3 in a cellular mobile radio system according to figure 1.

25 Figure 5 illustrates a base station in a cellular mobile radio system according to figure 1 with radio channels used according to figures 2 to 4.

30 Figure 6 illustrates a mobile station in a cellular mobile radio system according to figure 1 for communication with a base station according to figure 5 on control and digital communication channels according to figures 2 to 4.

Figure 7 illustrates actual error rate and estimation of error rate on a digital channel by using additional channel coder at the receiving side of a base or mobile station.

5 Figure 8 illustrates impulse response of digital channel radio link including transmission and receiving means and estimation of time dispersion on a digital channel using synchronization words.

DETAILED DESCRIPTION OF EMBODIMENTS

10 Figure 1 illustrates ten cells C1 to C10 in a cellular mobile radio system. Normally a method according to the invention is implemented in a cellular mobile radio system comprising many more cells than ten. However, for the purpose of explaining the invention ten cells may be sufficient.

15 For each cell C1 to C10 there is a base station B1 to B10 with the same number as the cell. Figure 1 illustrates base stations in the vicinity of cell centre and having omnidirectional antennas. The base stations of adjacent cells may however be colocated in the vicinity of cell borders and have directional antennas as is well known to those skilled in the art.

20 Figure 1 also illustrates ten mobile stations M1 to M10 within a cell and from one cell to another cell. Normally a method according to the invention is implemented in a cellular mobile radio systems comprising many more mobile stations than ten. In particular the mobile stations are normally times as many as the base station. However for the purpose of explaining the invention ten mobile stations may be sufficient.

25 Also illustrated in Figure 1 is a mobile switching centre MSC. The mobile switching centre illustrated in Figure 1 is connected to all ten illustrated base stations by cables. The mobile switching centre is connected by cables also to a fixed public switching telephone network or similar network with ISDN facilities. All cables from the mobile switching centre to base stations and cables to the fixed network are not illustrated.

5 In addition to the mobile switching centre illustrated there may be another mobile switching centre connected by cables to other base stations than those illustrated in Figure 1. Instead of cables other means may be used for base to mobile switching center communication, e.g. fixed radio links.

10 The cellular mobile radio system illustrated in Figure 1 comprises a plurality of radio channels for communication. The system is designated both for analogue information, e.g. speech, digitized analogue information, e.g. digitized speech, and pure digital information, e.g. pure data. In this application the term connection is used for a communication channel between a mobile station and another mobile station in the same system or another system or a fixed telephone or terminal in a fixed network connected to the cellular mobile radio system. Thus a connection 15 may be a call where two persons talk to each other but may also be a data communication channel where computers exchange data.

20 Figure 2 illustrates somewhat simplified a plurality of radio channels RCH₁ to RCH_{2n} within a frequency band. A first group of radio channels RCH₁ to RCH_N are used in the cellular mobile radio system for transmission of radio signals from base stations to mobile stations. A second group of radio channels RCH_{N+1} to RCH_{N2} are used in the cellular mobile radio system for transmission of 25 radio signals from mobile stations to base stations.

30 Some of the radio channels are used for control channels. Normally each base station has at least one control channel. Normally a control channel is not used for transfer of information on a connection but for monitoring and control of mobiles during setup of a connection, maintaining an established connection and handoff of an established connection. In Figure 3 is illustrated how the radio channel RCH_f all the time is used for a control channel CCH_K while the radio channel RCH_g all the time is used for a control channel CCH_R.

Some of the radio channels are used for analogue communication channels. Normally analogue communication are used for connections

where analogue informations is exchanged, e.g. telephone calls where two persons talk to each other. Normally one analogue communication channel is required for each such connection. When a radio channel is used for an analogue communication channel the information on the connection is transmitted with analogue modulated radio signals. In addition to the information on the connection the analogue communication channel may also be used for associated information, e.g. a supervisory audio tone. In figure 3 is illustrated how the radio channel RCHa all the time is used for an analogue communication channel ACHi while the radio channel RCHb all the time is used for an analogue communication channel ACHv. Normally each base station has at least one radio channel used for analogue communication channel.

Some of the radio channels are used for digital communication channels. Normally digital communication channels are used for connections where digital or digitized information is exchanged, e.g. data or digitized speech. The radio channels used for digital communication channels are divided into time slots and the time slots are grouped in frames. The time slots are allotted to digital communication channels whereby multiple digital channels share a common radio channel in time division multiplex. In Figure 3 is illustrated a radio channel RCHc having three time slots in each frame F. A first of the time slots is allotted to the digital communication channel DCH4, a second of the time slots is allotted to the digital communication channel DCH5 AND the third of the time slots is allotted to the digital communication channel DCH6. Thus the radio channel RCHc is used for three digital communication channels. Figure 3 also illustrates how the radio channel RCHd is used for three digital communication channels DCH7, DCH8 and DCH9 in a corresponding way.

In figure 3 the frames F of radio channels RCHc and RCHd have three time slots. Depending on the required bandwidth of the various digital communication channels it is conceivable to have less slots in a frame, e.g. two slots, or to have more slots in frame, e.g. six slots. When the digital communication channels are used for connections where digitized speech is exchanged, six time

slots may give too poor speech quality when the radio bandwidth is 30 kHz.

- On a radio channel used for digital communication channels the base or mobile station transmits a time slot identifier code with the radio signals at least in every time slot used for a connection.
- 5 On a particular radio channel, e.g. RCHc, the time slot identifier codes in different time slots are different. thus the time slots identifier code T11 is transmitted in the first time slot of radio channel RCHc allotted to digital communication channel DCH4. The time slot identifier code T12 is transmitted in the second time slot of radio channel RCHc allotted to digital communication channel DCH5. The time slot identifier code T13 is transmitted in the third time slot assigned to digital communication channel DCH6.
- 10
- 15 The same time slot identifier codes may be used on two or more radio channels, possibly all radio channels. Figure 3 illustrates how the time slot identifier T11 is transmitted in the first time slot of radio channel RCHd allotted to digital communication channel DCH7. The time slot identifier code T12 is transmitted in the second time slot of radio channel RCHd allotted to digital communication channel DCH8. The time slot identifier code T13 is transmitted in the third time slot of radio channel RCHd allotted to digital communication channel DCH9. Thus the time slot identifier code does not alone identify the channel but identifies
- 20
- 25 the time slot in a frame. It is conceivable to have one set of time slot identifiers T11 to T13 for channels with three slot frames F and have a different set of time slot identifiers T14 to T9 for six slot frames whereby the time slot identifier also may indicate the number of slots in a frame of the radio channel.
- 30 On a radio channel used for digital radio channels also transmitted a digital voice colour code with the radio signals at least in each time slot used for a connection. On a particular radio channel the same digital voice colour code is transmitted with the radio signals in different time slots. Figure 3 illustrates
- 35 the transmission of the same digital voice colour code VC1 in all

time slots of the radio channel RCHc. Normally the same digital voice colour code is used for all radio channels to and from a particular base station, e.g. the digital voice colour code VC1 is used for all radio channels to and from base station BS1.

- 5 Some adjacent base stations may use the same digital voice colour code e.g. base stations B2, B6 and B7 may use the same digital voice colour code as long as base station BS1. Other base stations use a different digital voice colour code, e.g. base stations B4, B5, B9 and B10 all use the digital colour code VC4. Still other
10 base stations use another different digital voice colour code, e.g. base stations B3 and B8 use the digital voice colour code VC7.

The radio signals on a radio channel used for time division multiplex digital communication channels are transmitted in bursts. Figure 4 illustrates a burst in a time slot separated by guard spaces from the end of the preceeding burst and the beginning of a succeeding burst in adjacent time slots.
15

A transmitted burst comprises at least a time slot identifier, abbreviated T1, and a digital voice colour code, abbreviated VC, and normally also information to be forwarded to the other party of the connection as well as connection or channel associated information for control or monitoring purposes.
20

It is well known to those skilled in the art that there is a need for receiver synchronisation in time division multiple access radio communication systems. For this purpose it is well known to transmit synchronisation words or patterns in each burst and to transmit particular frame synchronisation words or patterns from a master or base station to a slave or mobile station. Preferably the time slot identifier code according to the present invention may also be used for synchronisation of receiver to transmitter. Selecting time slot identifier codes in this way means that the implementation of time slot identifier codes according to the present invention does not require any dedicated bits for the time slot identifier. Teorectically there exist a number of uncorre-
25
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lated binary multi bit words, e.g. 26 bit words, which may be used as different synchronisation words. According to the present invention one unique time slot identifier and synchronisation words is required for each time slot in a frame on the channel. For the purpose of both time slot identification and synchronisation the time slot identifier code words are to be chosen for minimum correlation between each other and between themselves except for correlation to themselves in phase. Thus a first time slot identifier code used to identify the first time slot of a frame according to the present invention and to synchronise receiver to transmitter shall exhibit low correlation to an identical code when substantially out of phase but exhibit high correlation to an identical code when substantially in phase. A second time slot identifier code used to identify the second time slot of a frame and to synchronise receiver to transmitter shall also be exhibit low correlation to an identical code when out of phase but exhibit high correlation to an identical code when substantially in phase. All other time slot identifiers used on the channel should also exhibit low correlation to an identical code when out of phase but exhibit high correlation to an identical code when in phase. Any time slot identifier code used on the channel should also exhibit low correlation to any other time slot identifier code used on the channel irrespective of phase relationship, i.e. of in phase or out of phase. Once the requirements on time slot identifier codes are given one skilled in the art will be able to select proper time identifier codes without inventive activity. However, for the convenience of those not skilled in the art the following eight 26 bit words are given as examples of possible time slot identifier codes for an eight slot frame:

code for time slot 1:

(0,0,1,0,0,1,0,1,1,0,0,0,0,1,0,0,0,1,0,0,1,1,1,1,)

code for time slot 2:

(0,0,1,0,1,1,0,1,1,1,0,1,1,1,1,0,0,0,1,0,1,1,0,1,1,1,1,)

code for time slot 3:

(0,1,0,0,0,0,1,1,1,0,1,1,1,0,1,0,0,1,0,0,0,0,1,1,1,0,)

code for time slot 4:

(0,1,0,0,0,1,1,1,1,0,1,1,0,1,0,0,0,1,0,0,0,1,1,1,1,0,)

code for time slot 5:

(0,0,0,1,1,0,1,1,1,0,0,0,0,0,0,1,1,0,1,0,1,1,)

code for time slot 6:

(0,1,0,0,1,1,1,0,1,0,1,1,0,0,0,0,0,1,0,0,1,1,1,0,1,0,)

code for time slot 7:

5 (1,0,1,0,0,1,1,1,1,0,1,1,0,0,0,1,0,1,0,0,1,1,1,1,1,1,)

code for time slot 8:

(1,1,1,0,1,1,1,1,0,0,0,1,0,0,1,1,1,0,1,1,1,0,0,)

When there are less than eight time slots in a frame on the radio channel less than all of the given code words may be used. However 10 there may be advantages in using other than the given code when there are only three time slots in a frame.

It of course conceivable to use binary time slot identifier codes having more or less than 26 bits. Longer code words offer some 15 advantages but has the disadvantage of occupying more of the available space in a burst.

In the advanced mobile phone service system, abbreviated AMPS, there is a supervisory audio tone, abbreviated SAT, transmitted e.g. on analog communication channels. The reason for transmitting the SAT in AMPS is that in an interference limited mobile radio 20 communication network there should be some means for the receiving entity (e.g. base station) to identify the transmitting entity (e.g. mobile station) or at least with high likelihood exclude interchange of transmitter entity without the need for continuous transmission of transmitter identity. The object of the digital 25 channel code in a cellular mobile radio system according to the present invention is at least partly the same as that of SAT in AMPS. The number of different SATs is three in AMPS. The number of different digital voice colour codes in a cellular mobile radio system according to the present invention is preferably much 30 higher than three, mainly to support a proper introduction of discontinuous transmission. The number of different digital colour codes should be high enough to guarantee that cells where the same radio channels are used having the same colour code ar at a geographical distance from each other sufficient not to cause co- 35 channel interference. In case discontinous transmission is used

the disturbing signal of an other station should not significantly exceed the level of noise. For this purpose an eleven bit colour code seems quite sufficient in normal cases. However, the longer color codes the longer connection setup and handoff procedures.

5 Considering the possible load on control channels eight bit digital voice codes seems to be a good compromise. Preferably the digital voice channel code is binary word having eight bits, enabling theoretically 256 different voice colour codes.

A separate digital voice colour code requires space in a burst thus reducing the space available for speech or data on the connection. However, the following implementation of voice code does not require any dedicated bits for the digital voice code in a burst on a digital communication channel.

10 At the transmitting side the digital voice colour code is added bit by bit modulo two to a field under error detection but without error correction within the information part of the burst after channel coding. On the receiving side the bursts in the time slots are deinterleaved and checked. This check is done by adding bit by bit the known digital voice colour code modulo two, as is done at 15 the transmitting side, before channel decoding and error detection. If no error is found after adding the digital voice colour code modulo two the information part of the burst was sent from the expected transmitter and not from an interferer.

20 Figure 5 illustrates a base station in a cellular mobile radio system according to figure 1 with radio channels according to figures 2 to 4. The base station is designed for transmission and reception on a plurality of radio channels used for digital communication channels, analog communication channels and control channels. In figure 5 not all base station equipment for all 25 channels is illustrated. Normally a base station has equipment for more channels, in particular analogue communication channels, as well as other equipment for power supply, maintenance etc but the illustrated equipment is believed sufficient for the understanding 30 of the present invention.

- The base station is connected to a mobile switching centre by six trunks. A first incoming trunk for digital communication channels is connected to a digital trunk demultiplexor and interface DMU-D. A second incoming trunk for analog communication channels is connected to an analog trunk interface and demultiplexor DMU-A. A third incoming trunk for control channels and base station control information is connected to a trunk interface and control information demultiplexor DMU-C. A first outgoing trunk for digital communication channels is connected to a digital multiplexor and trunk interface MUX-D. A second outgoing trunk for analog communication channels is connected to an analog multiplexor and trunk interface MUX-A. A third outgoing trunk for control channels and base station information is connected to a control information multiplexor and trunk interface MUX-C.
- For each of the outgoing radio channels used for digital communication channels the base station comprises digital analog channel coding means, DCC, connected to a digital trunk demultiplexor and interface, DMU-D, modulo two adding means, M2A, burst generation means, BG, modulator means, MOD, and power amplification means, PA, connected to an antenna. Two such outgoing radio channels share a digital voice colour code means, VCS, connected to their modulo two adding means. Two such outgoing radio channels also share a time slot identifier code means, TIS, connected to their burst generators.
- For each of the outgoing radio channels used for analog communication channels the base station comprises analog transmission channel processing means, ATC, connected to an analog trunk interface and demultiplexor, DMU-A, modulator means, MOD, and power amplification means, PA, connected to an antenna.
- For each of the outgoing radio channels used for control channels the base station comprises outgoing control channel processing means, CTC, connected to the trunk interface and control information demultiplexor, DMU-C, modulator means, MOD, and power amplification means, PA, connected to an antenna.

For each incoming radio channel used for digital communication channels the base station comprises radio receiver means, REC, connected to an antenna, radio signal strength or level measuring means, SLM, analog to digital conversion means, A/D, multipath equalizer and burst synchronizing and time slot recognition and automatic frequency control means, EQ-AFC, modulo two adding means, M2A and digital channel decoder means, DCD, connected to a digital multiplexor and trunk interface, MUX-D.

5 Two incoming radio channels used for digital communication share digital voice colour code means, VCS, connected to their modulo two adding means. Two such incoming radio channels also share digital channel bit error measuring means, BEM, connected to their digital channel decoders, DCD.

10 15 For each incoming radio channel used for analog communication channels the base station comprises radio receiver means, REC, connected to an antenna, radio signal strength or level measuring means, SLM, and incoming analog channel processing means, ARC, connected to an analog multiplexor and trunk interface, MUX-A.

20 For each incoming radio channel used for control channel the base station comprises radio receiver means, REC, connected to an antenna, radio signal strength or level measuring means, SLM, and incoming control channel processing means, CRC, connected to the control information multiplexor and trunk interface, MUX-C.

25 30 All modulation means and radio receiver means are connected to frequency synthesizer means, FS. The frequency synthesizer means are controlled by a central processor, CP. The CP also controls the DCCs, VCSs, BGs, EQ-AFCs, DCDs, BEM, ATC, ARC, CTC, CRC and MUX-C. Preferably the central processor is not the only processor in the base station but other means may also comprise processors, in particular the ATC, ARC, CTC, CRC and EQ-AFCs.

The base station according to Figure 5 is intended for communication with mobile station only having equipment for analog communication channels and control channels. The base station is

also intended for communication with mobile stations only having equipment for digital communication channels and control channels. The base station is also intended for communication with dual mode mobile stations designed for communication on both analog and 5 digital communication channels as well as channels.

Mobile stations designed only for analog communication channels may be of a kind well known to those skilled in the art and operate according to AMPS standard. Furthermore, a method according to the invention can only be implemented with mobile stations with means 10 for communication both on digital and on analogue radio channels. Accordingly there is no need for disclosing a mobile designed only for analogue radio channels or its operation here. Neither is there any need for describing parts of a dual-mode mobile station used only for communication on analogue radio channels and control 15 channels.

Figure 6 illustrates parts of a dual-mode mobile station in a cellular radio system according to Figure 1 for communication with a base station according to figure 5 on radio channels according to figures 2 to 4. The parts illustrated are involved in communication on digital radio channels. A dual mode mobile station capable of using both analog and digital radio channels thus includes in addition to the means illustrated in figure 6 first 20 analog signal processing means connected to the microphone and the modulator and second analog signal processing means connected to the radio receiver IF stage and to the loudspeaker. The first and second analog signal processing means are merely indicated by one 25 block and are controlled by the microprocessor.

The mobile station comprises a microphone connected to analogue to 30 digital speech coding means for coding speech or sound into a binary code with a bitrate or bandwidth less than 11 kHz. preferably about 7 to 8 kHz. Connected to the speech coding means is channel coding means for interleaved error protecting coding of the digital information from the speech coder. The channel coder is connected to a modulo two adder for modulo adding of a digital 35 voice colour code to the digital information from the channel

coder. The modulo two adding means are connected to a burst generator for collecting information to be transmitted in a burst and arranging the information including time slot identifier code into a proper burst. When the mobile is used for data transmission
5 on a connection or during connection setup the burst generator puts data and/or control information in the burst instead of digitized speech. Such information may be supplied from a keyboard via a microprocessor and the channel coder or directly from the microprocessor. A modulator is connected to the burst generator
10 for receiving digital information to be modulated on a radio frequency carrier from a frequency synthesizer and be amplified in a power amplifier. The modulation method used may be continuous phase modulation or other types suitable for digital information transmission. The power amplifier is connected to an antenna via
15 a duplex and is controlled from the microprocessor.

The mobile station also comprises a radio receiver connected to the duplexer, a radio signal or level means and analog to digital conversion means. The radio receiver comprises RF and IF stages with filter, demodulator etc. Means for equalizing the digital communication channel and automatic frequency control and automatic gain control are connected to the radio receiver and the input of a modulo two adder. The modulo two adder adds a digital voice colour code modulo two to the digital information from the equalizer. The modulo two adder output is connected to a channel decoder for deinterleaving and error detection and correction of digital information from the modulo two adder. Means for converting digital information to analogue information or speech are
20 connected to the channel decoder and a loudspeaker.
25

When the mobile station is tuned to a radio channel used by a base station for control channel some of the illustrated equipment of the mobile station is not used, in particular not channel and speech decoder. When control and monitoring information is transmitted from the base station on the control channel according to the AMPS standard the microprocessor receives and interprets
30 signals from the analog to digital converter.
35

Except for the time slot identifier and the digital voice colour code and means for introducing, recognizing and removing them from the flow of information the mobile radio according to Figure 6 may operate on time division multiplex digital communication channels in a way similar to known digital mobile radio stations, e.g. of the kind disclosed in the GSM standard or in Ericsson Review No. 3, 1987.

Accordingly there is no need here to further describe in detail the overall operation or the detailed operation of the various means. Regarding the time slot identifier code and the digital colour code the mobile station comprises means for storing all possible time slot identifier codes and digital voice colour codes to be used in communication with a base station. The microprocessor receives instructions from the base as to which codes to use in the particular connection and reads the codes from the stores and supplies the modulo two adders and burst generator with appropriate time slot identifier code and digital voice colour code.

When receiving radio signals from a base station synchronisation and recognition of time slot identifier is performed in the equalizer in cooperation with the microprocessor. Measurement of bit error rate on an established connection is performed in the channel decoder in cooperation with the microprocessor. Equalizers and methods of synchronisation and bit error measurement are well known to those skilled in the art. Accordingly there is no need here to further describe such methods or means for performing them. However for the person not skilled in the art a brief explanation on bit error measurement for digital radio channels will be given in connection with figure 6 and 7.

In this example it is assumed that at least a part of the digital information transmitted on the digital radio channel is protected by an error correction code. As an example, both the GSM and TIA systems use 20 ms speech blocks with part of the speech coder output bits protected by an error correction code. Since the speech coder operate block wise so does the error correction applied.

The actual bit error rate, BER, of the radio channel including certain transmitting and receiving means at base and mobile station is of course due to the difference between the information actually leaving the channel coder of the transmitter and the information received by the receiver decoder. This is illustrated in the left hand part of figure 7. An estimate of the actual bit error rate can be done by reencoding the decoded data in the receiver and compare this bit stream with the input to the channel decoder of the receiver. This is illustrated in a principal way in the right hand part of figure 7. For the purpose of this an additional channel encoder may be used and in figure 6 an additional encoder has been incorporated in the receiving part of the mobile station. Corresponding additional encoder means may of course be incorporated in the receiving part of a base station according to figure 5. For comparing corresponding digital symbols according to figure 7 the central processor in the base station and the microprocessor of the mobile station may be used.

If the channel decoder in the receiver has corrected all bit errors in a block of n bits then the bit by bit comparison of the received data with the reencoded data will be equal to the actual number of bit errors in the block of n bits. The number of bit errors is the number of different bit values in the above mentioned comparison. Thus, in this case the estimated number of bit errors is equal to the true value of bit errors. The bit error rate in the block of n bits is then the number of bit differences divided by n .

If the channel decoder of the receiver is not capable of reproducing the transmitted n bits e.g. the channel decoder makes an error, the estimated bit error rate will not be equal to the actual error rate. Nevertheless, for bit error rates in the range of interest the estimate will be highly correlated to the actual rate. The difference between the estimated and actual rate can be seen as a measurement noise term. The system is intended to operate with a radio link quality where most of the blocks can be corrected by the channel decoder, otherwise the audio quality would be unintelligible, thus the measurement noise is limited. If

the radio link quality is so low that almost every block is uncorrectable by the channel decoder, the output from the decoder is more or less a random pattern with low correspondence with the received block. This can for instant happen if the amount of time dispersion is much more than the equalizer can handle. The important thing is that the bit error estimate will still cause a generally correct conclusion, i.e. too low radio channel quality, because the difference between two blocks with almost no correspondence will result in that, in average, every second bit comparison will indicate a bit error. Thus the bit error rate will be estimated to be 50%.

In summary, independent of the radio link quality this method described will indicate to the receiver the quality of the radio link in terms of a bit error estimate.

The procedure for connection setup in a cellular mobile radio system according to figure 1 with a base station according to figure 5 and a mobile station according to figure 6 may be similar to the corresponding procedure in AMPS when the channel to be used is an analogue communication channel. However, when the channel to be used for the connection is a digital communication channel according to figures 3 and 4 the base station informs the mobile station not only on radio channel but also on time slot and digital voice colour code to be used. During the setup procedure the base station then also informs the mobile station on a plurality of radio channels the signal strength on which to be measured by the mobile. Normally this plurality of radio channels are the radio channels used for channels by adjacent bases/cells. Depending upon the movement of the mobiles as well as other circumstances a new plurality of radio channels may be selected and corresponding information be transmitted to the mobile from the responsible base station during the course of the connection. During the course of a connection for which a digital communication channel is used the mobile measures the signal strength of signals on the given plurality of radio channels. Measurements may be done during time slots not used by the digital communication channel. The mobile station also measures signal strength on the digital communication

channel used for the established connection and the bit error rate on the established connection. The mobile station transmits results of its measurements, preferably averaged, frequently to the base station, preferably twice a second.

- 5 In addition to or instead of measuring signal strength and bit error rate the mobile station may estimate time dispersion on the used digital radio channel using the time slot identifier code words and the equalizer in a way to be described later. The mobile station may transmit the estimation in addition to or instead of signal level or bit error rate measurement results frequently to the base station.

The base station also measures signal strength on the digital communication channel used for established connection and the bit error rate on the established connection. The base station may 15 also estimate time dispersion on incoming radio channels used for digital communications channels. This may be done using the time slot identifier codes and equalizer in a way to be described later. The base station processes and analyzes the results of its own measurements and/or time dispersion estimates and the measurement and/or time dispersion estimates of the mobile and compares 20 with criteria for handoff. When according to the results and criteria a handoff to another base station is desirous the base station informs the mobile switching centre indicating at least one target base station assumed suitable for taking over the 25 responsibility for the communication with the mobile.

The mobile switching centre request the target base station(s) to measure signal strength on a radio channel in a time slot used by the mobile for the established connection. The mobile switching centre also informs the target base station on the digital colour 30 code used by the mobile station.

The target base station(s) tune (s) a receiver to the radio channel indicated by the mobile switching centre and uses the time slot identifier of the indicated time slot for burst synchronisation. The target base station checks the appearance of the digital

colour code indicated by the mobile switching centre and measures the signal strength of the burst signals provided the digital colour code is correct. The target base station then transmits its results of signal strength measurement to the mobile switching centre. The target base station also informs the mobile switching centre on the result of the checking of the appearance of the digital colour code, that is whether the digital voice colour code appeared in the burst in the time slot of the radio channel.

The mobile switching centre determines whether handoff to a target base should be performed taking the results of signal strength measurements of target base(s) into account as well as other circumstances, e.g. traffic load.

When the mobile switching centre determines that handoff to an other base station and digital radio channel shall be performed it transmits to responsible base and target information on new radio channel, new time slot and new voice colour code to be used by the mobile station for the connection after handoff and new radio channel to be used by target base station for the connection after handoff.

The responsible base station forwards information about the two new radio channels, new time slot and new digital colour code to the mobile. After receiving this information the mobile station tunes to the new radio channel to be used for the connection by the target base station and looks for the new time slot identifier code in received signals on the radio channel. The mobile station uses the new time slot identifier code in received signals for burst synchronisation. After synchronisation and tuning its transmitter to the new radio channel the mobile station begins transmitting bursts in the new time slot on the new radio channel. The new digital colour code is transmitted with each burst.

The target base station tunes a receiver to the new channel to be used for the connection by the mobile station and looks for the new time slot identifier code. The target base station uses the time slot identifier code for synchronisation. The target base

station then looks for the new digital colour code in signals in the new time slot of the new channel. If the target base station identifies the new digital colour code in the bursts in the new time slot of the new radio channel this is reported to the mobile switching centre. The mobile switching centre then interpretes the handoff as sucessful and acts accordingly. After sucessful handoff the former target base station now being responsible base station informs the particular mobile station on a new plurality of radio channels the signal strength on which to be measured by the mobile.

In the embodiment of the handoff method described above the responsible base station and mobile station use the same time slot identifier code and the same digital voice colour code. However, it is conceivable to use different time slot identifier at base and mobile stations for a particular connection.

In the embodiment of the handoff method described the mobile station measures signal strength on radio channels used for control by base station. However, it is conceivable to request mobile station to measure signal strength on radio channels used for digital communication channels by base station, in particular when there are no radio channels entirely used for control channels by base stations.

The procedure for intracell handoff from a digital radio channel to an analog radio channel of the same base station is of course less complex than the described handoff procedure because responsible base station and target base station is the same. When according to the results and criteria a handoff from a digital radio channel to an analog radio channel is desirous the base station informs the mobile switching centre. After approval by the mobile switching centre the base station may perform the normal procedure of the system for intracell handoff to a new analog radio channel, the procedure comprising transmitting informing to the mobile station on the new radio channel to be used. If the invention is implemented in a mobile radio system where handoff between analog channels is performed according to the AMPS

standard the this procedure may also be used for handoff from digital to analog handoff due to time dispersion. If the invention is implemented in a mobile radio system where handoff between analog channels is performed according to an other standard then 5 this procedure may also be used for handoff from digital to analog channel due to time dispersion. Accordingly there is no need for describing a complete such a procedure here. Suffice it to say that there is no transmission of information on time slot identifier code or digital voice colour codes from base to mobile but 10 instead information on supervisory audio tones or similar means may be transmitted from base to mobile.

The procedure for handoff from a digital channel of one base station to an analog channel of an other base station is of course a little more complex than the intracell handoff procedure because 15 more than one base station is involved. However, the normal procedure of the system for handoff to a new base station and a new analogue channel may be used, the procedure comprising transmitting information to the mobile station on new base station and new analog channel. If the invention is implemented in a mobile radio 20 system with analog channel handover according to the AMPS standard then this procedure may be used. If the invention is implemented in a system with different handover procedure then this handoff procedure is well known to those skilled in the art. Anyone not skilled in the art of procedures for analog channel handover is 25 recommended to study the specifications on the NMT and AMPS systems or the specification of the system where the invention is to be implemented and possibly also published patents in the field of handover. Accordingly there is no need for describing such a handover procedure here.

30 The intelligence and decision making in a mobile cellular system according to figure 1 may be more or less centralized to the mobile switching centre or more or less decentralized to the base stations. In a decentralized system more or less of the functions 35 of the mobile switching office during the handoff preparation and performance may instead be performed in the responsible and/or target base stations.

A method of directly estimating time dispersion and determining when the time dispersion is too big will now be described in connection with figures 6 and 8.

The signals from the A/D converter in figure 6 are sampled a number, j , of times per bit time. In the equalizer the received signals are correlated with the locally stored time slot identifier. In this way an estimate $Y(t)$ of the channel impulse response is calculated in the equalizer and made available for processing and equalization in the central processor. In figure 8 the absolute values of $Y(t)$ are indicated as a function of time. The distance between adjacent indicated values of $Y(t)$ is the symbol (bit) time, T , divided by j . A process of estimating the impulse response is described in detail in the earlier cited article "Bit synchronization and timing sensitivity in adaptive Viterbi Equalizers for Narrowband-TDMA Mobile Radio Systems". Accordingly for a further understanding of figure 8 reference is made to this article.

The equalizer correlates the received signal with the time slot identifier TSID over a movable time span W_t , hereafter called the measurement window, which is indicated in figure 8. The length of the measurement window represent the amount of time dispersion that the equalizer can handle e.g. energy within the measurement window contributes in a positive manner for the demodulation/bit detection process whereas energy outside this window act as an interfering signal. The length of the gliding window is a design parameter in the construction of the equalizer. With a long window the equalizer can handle more time dispersion but the complexity and power consumption will increase.

The equalizer moves the gliding window in relation to the impulse response, calculates the impulse energy inside and places the window in such a position in relation to the impulse response that the part of the energy, c , inside the window is maximized. This part of the impulse inside the window is shadowed in figure 8. Maximizing the part of the energy inside means that the impulse energy, r , outside the measurement window is minimized. The

positioning of the movable window is further elaborated in the above mentioned reference. The estimation of the impulse response is an inherent feature of the equalizer. Depending on the equalization algorithm used, the impulse response may be estimated 5 only once in each burst, as described above, but may also be updated during the duration time of the burst. Thus, the central processor may obtain at least once every burst the amount of useful energy, c , and the interfering energy, r , from the equalizer.

10 The receiver processor can now average the individual c and r values over a plurality, n , of bursts so that influence of estimation errors, short time variations of radio propagation properties etc is reduced. The averaged quantities of c and r are denoted C and R respectively. Since C represents useful energy and R represents disturbing energy the C/R ratio thus represents a 15 figure of merit of the ability of the equalization to reproduce the transmitted data in a way similar to the well known S/N ratio used for telecommunication in general.

If the C/R ratio is low the bit error rate will be high. An 20 criteria that can be used to determine if the time dispersion is too severe to enable reproduction of an audio signal or data transmitted is to make the processor comparing the C/R with a threshold, K . Thus if C/R is less than K , the processor has determined that the time dispersion is so big that problem with 25 the time dispersion currently exists.

Of course the described method of estimation of time dispersion on a digital radio channel can be done either at the base station or at the mobile station or at both. The invention is not limited to the use of this particular method but decisions to handover to an 30 other channel may be based on other methods of estimation of time dispersion.

CLAIMS

1. A method in a mobile radio system, having both analog and digital radio channels, for maintaining a connection having started on a digital radio channel, when the dispersion on the used digital radio channel tends to exceed a maximum dispersion characterized in performing handover to an analog radio channel of the system.
2. A method according to claim 1, characterized in estimating time dispersion on a used digital radio channel indirectly by measuring both symbol error rate and signal level on the digital radio channel.
3. A method according to claim 1 characterized in estimation at a mobile station of time dispersion of a digital radio channel used by a base station for transmission to the mobile station, and reporting estimations from the mobile station to the base station.
4. A method according to claim 1 characterized in estimation at a base station of time dispersion of a digital radio channel used by a mobile station for transmission of information to the base station.
5. A method of maintaining a connection according to claim 1 in a multi channel mobile radio system comprising base stations and mobile stations designed both for digital transmission of information using digitally modulated radio signals on a digital radio channel and for analog transmission of information using analog-modulated radio signals on an analog radio channel, which mobile stations comprise equalizers designed for a particular maximum time dispersion of digitally modulated radio signals, characterized in estimation of the time dispersion of at least the digital radio channel used by a mobile radio station for the connection, and when the time dispersion of the digital radio channel is estimated to be too big performing handover to an analog radio channel of the system assumed to have sufficiently

good radio propagation properties if such an analog radio channel is available to the mobile station.

6. A method according to claim 5, characterized in estimation of the time dispersion of at least one possible digital radio channel available to the mobile station during the time the connection continues on the analog radio channel, and when the time dispersion of a digital radio channel available to the mobile station is estimated to be sufficiently small performing handover from the analog radio channel to this available digital radio channel.

10 7. A method of maintaining a connection according to claim 1 in a multi channel mobile radio system comprising base stations and mobile stations designed both for digital transmission of information using digitally modulated radio signals on a digital radio channel and for analog transmission of information using analog-modulated radio signals on an analog radio channel, which mobile stations comprise equalizers designed for a particular maximum time dispersion of digitally modulated radio signals, characterized in
estimation of the radio propagation properties of at least a first
20 digital radio channel used by a mobile radio station for an ongoing connection, estimation of the radio propagation properties of at least a second possible digital radio channel to the mobile station at least when the radio propagation properties of the first digital radio channel are estimated to be too poor, performing handover to the second digital radio channel when the radio propagation properties of the first radio channel are estimated to be too poor while the radio propagation properties of the second radio channel are estimated to be sufficiently good, and in the absence of any other available digital radio channel to the mobile
25 station having sufficiently good radio propagation properties when the radio propagation properties of the first radio channel are estimated to be too poor performing handover to an analog radio channel estimated to have sufficiently good radio propagation properties if such an analog radio channel is available to the
30 mobile station.
35

8. A method according to claim 7, characterized in estimation of the radio propagation properties of at least one possible digital radio channel to the mobile station during the time the connection continues on the analog radio channel, and when the radio propagation properties of a digital radio channel available to the mobile station is estimated to be sufficiently good performing handover from the analog radio channel to this available digital radio channel.

9. A method of maintaining a connection according to claim 1 in a multi channel mobile radio system comprising base stations and mobile stations designed both for digital transmission of information using digitally modulated radio signals on a digital radio channel and for analog transmission of information using analog-modulated radio signals on an analog radio channel, which mobile stations comprise equalizers designed for a particular maximum time dispersion of digitally modulated radio signals, characterized in

estimation at a mobile station having an ongoing connection on a digital radio channel of the signal level and symbol error rate of the digital radio channel used by that mobile radio station for the ongoing connection, estimation of the radio propagation properties of at least a second possible digital radio channel to the mobile station at least when the symbol error rate of the used digital radio channel tends to be too high without the signal level on the used digital radio channel tending to be too low, performing handover to the second digital radio channel when the symbol error rate of the used radio channel is estimated to be too high while the radio propagation properties of the second radio channel are estimated to be sufficiently good, and in the absence of any other available digital radio channel to the mobile station having radio propagation properties estimated to be sufficiently good when the symbol error rate of the used digital radio channel is estimated to be sufficiently high performing handover to an analog radio channel estimated to have sufficiently good radio propagation properties if such an analog radio channel is available to the mobile station.

10. A method according to claim 9, characterized in estimation of the radio propagation properties of at least one possible digital radio channel to the mobile station during the time the connection continues on the analog radio channel after handover, and when the 5 radio propagation properties of a digital radio channel available to the mobile station is estimated to be sufficiently good performing handover to this available digital radio channel.

11. A method of maintaining a connection according to claim 1 in a multi channel mobile radio system comprising base stations and 10 mobile stations designed both for digital transmission of information using digitally modulated radio signals on any of a plurality of digital radio channels and for analog transmission of information using analog-modulated radio signals on any of a plurality of analog radio channels, which mobile stations comprise equalizers designed for a particular maximum time dispersion of 15 digitally modulated radio signals, characterized in estimation at a mobile station having an ongoing connection on a digital radio channel of the signal level and symbol error rate of both the digital radio channel used by that mobile radio station for the ongoing connection and for at least one other possible digital 20 radio channel to the mobile station, estimation of the radio propagation properties of at least a second possible digital radio channel to the mobile station at least when the symbol error rate of the used digital radio channel tends to be too high without the 25 signal level on the used digital radio channel tending to be too low, performing handover to one other digital radio channel when the symbol error rate of the used radio channel is estimated to exceed a first value while the signal level of the other digital radio channel is estimated to exceed a second value and the symbol 30 error rate of the other radio channel is estimated to be lower than a third value, and in the absence of any other available digital radio channel to the mobile station having a sufficiently high estimated signal level and sufficiently low symbol error rate when the symbol error rate of the used digital radio channel is 35 estimated to exceed the first value performing handover to an analog radio channel estimated to have sufficiently high signal

level if such an analog radio channel is available to the mobile station.

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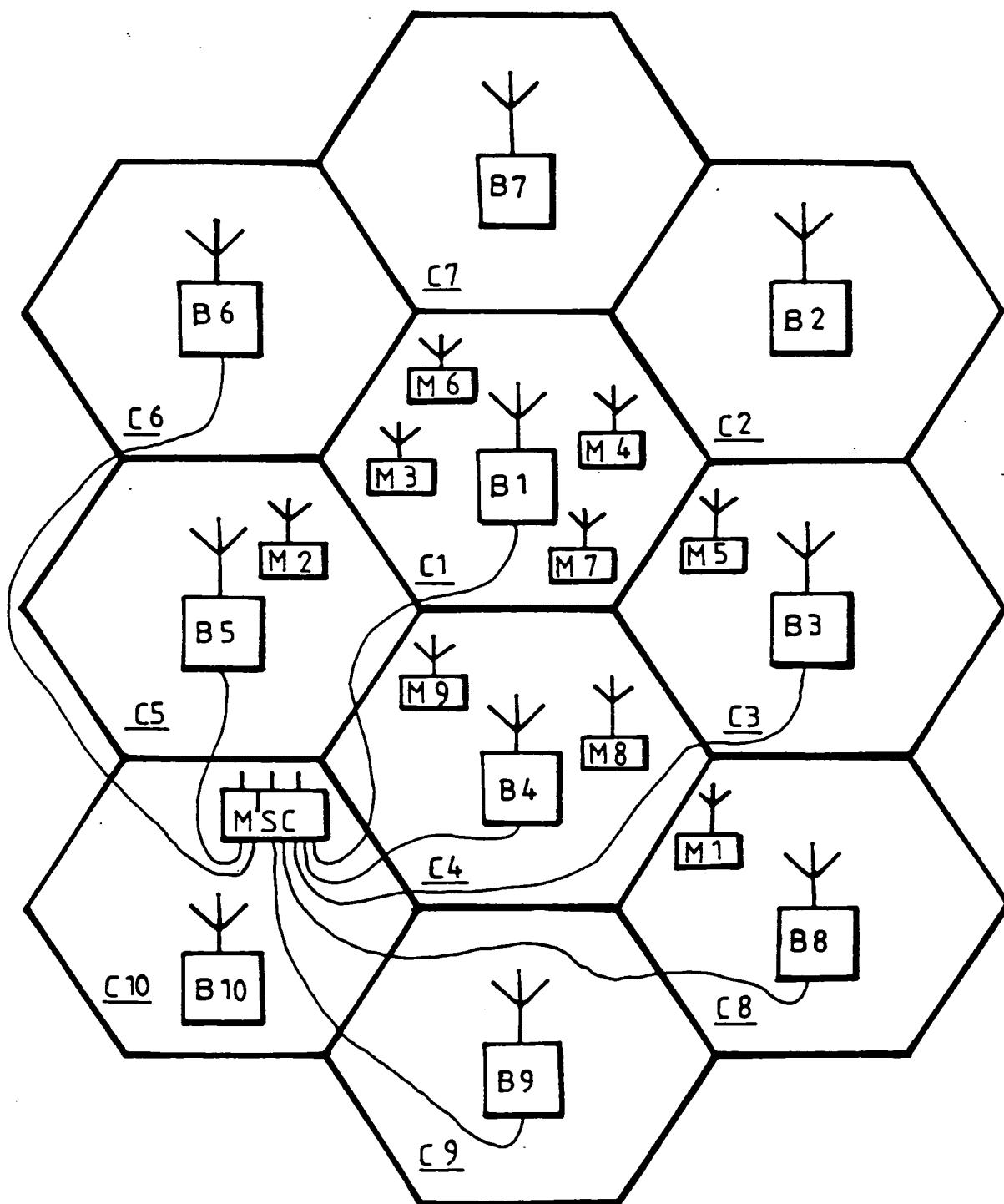


Fig.1

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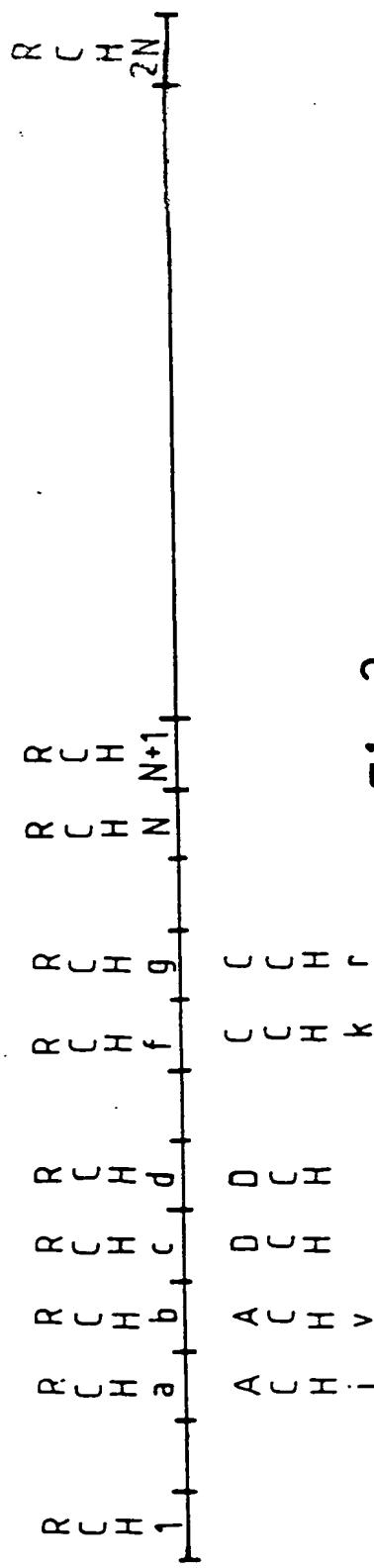


Fig.2

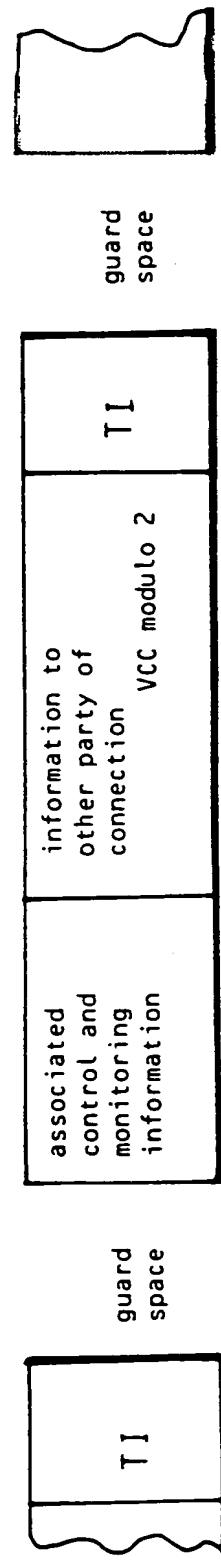


Fig.4

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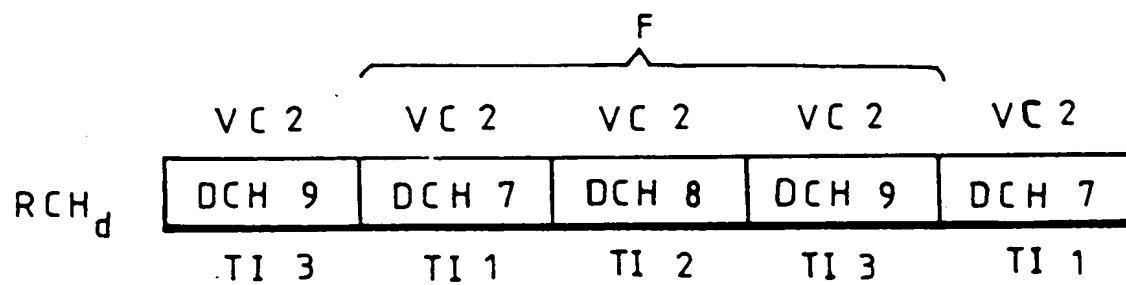
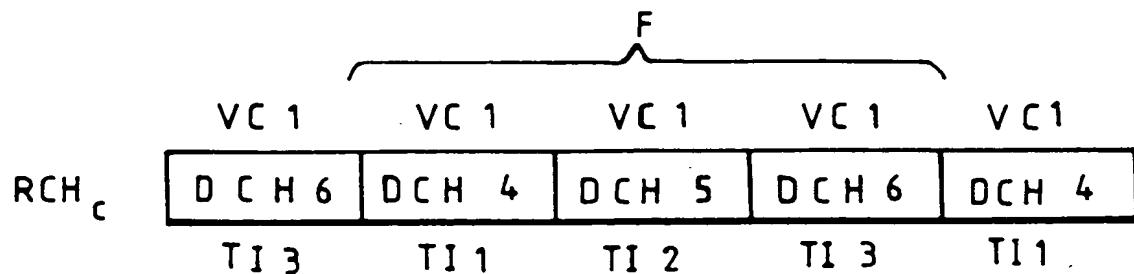
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RCH_a

ACHi

RCH_b

ACHv

RCH_f

CC Hk

RCH_g

CC Hr

Fig.3

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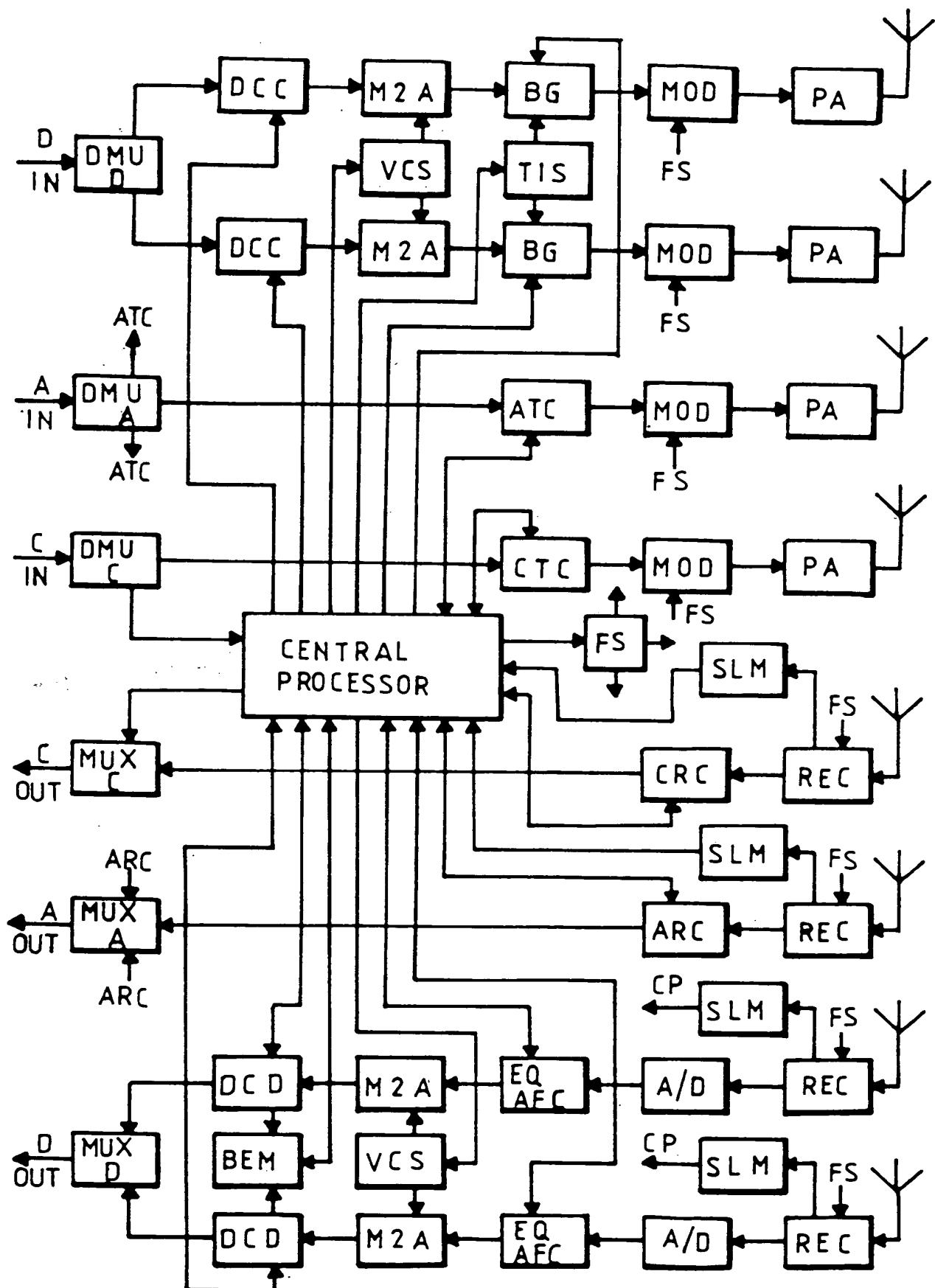


Fig.5

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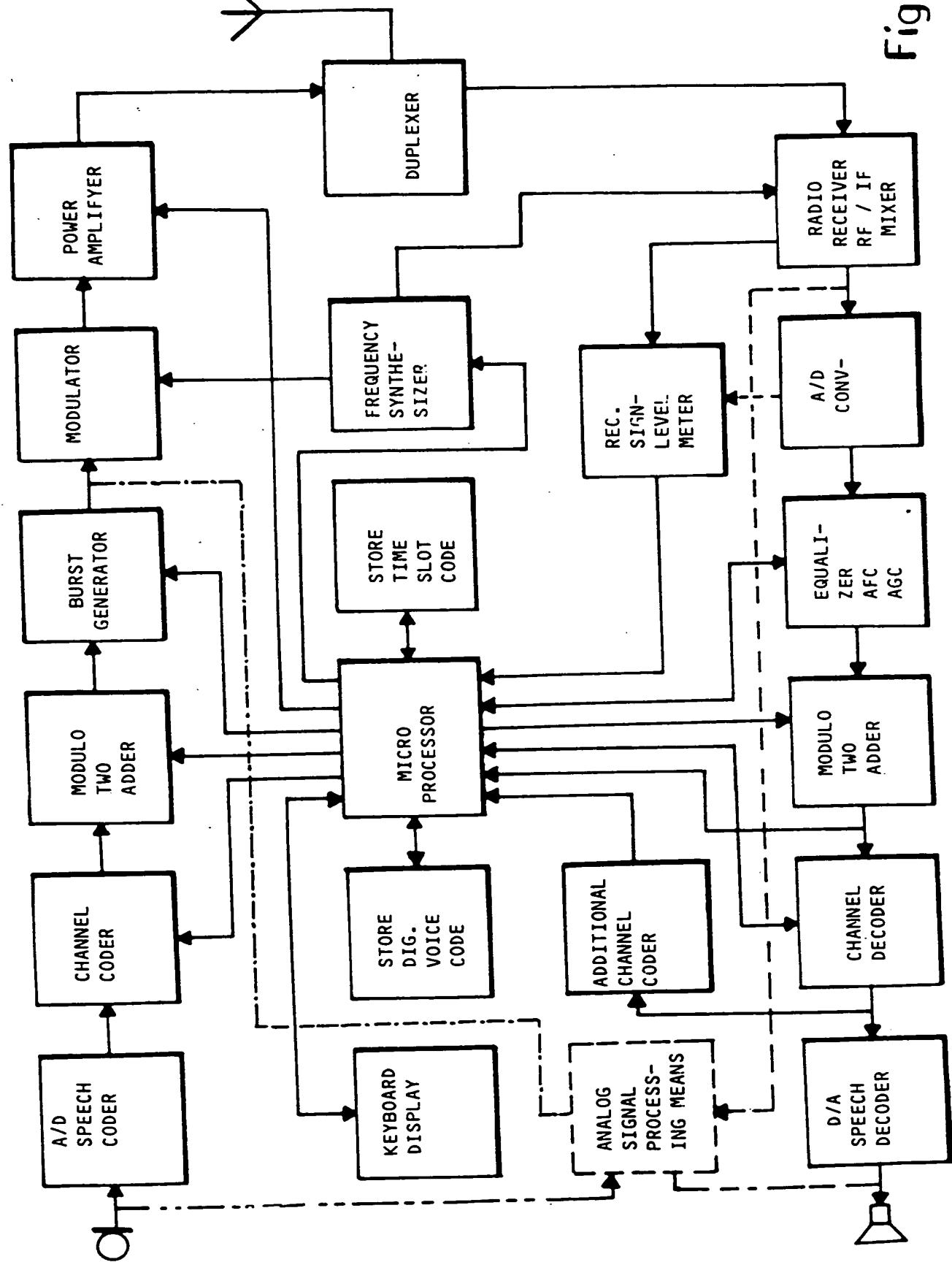


Fig 6

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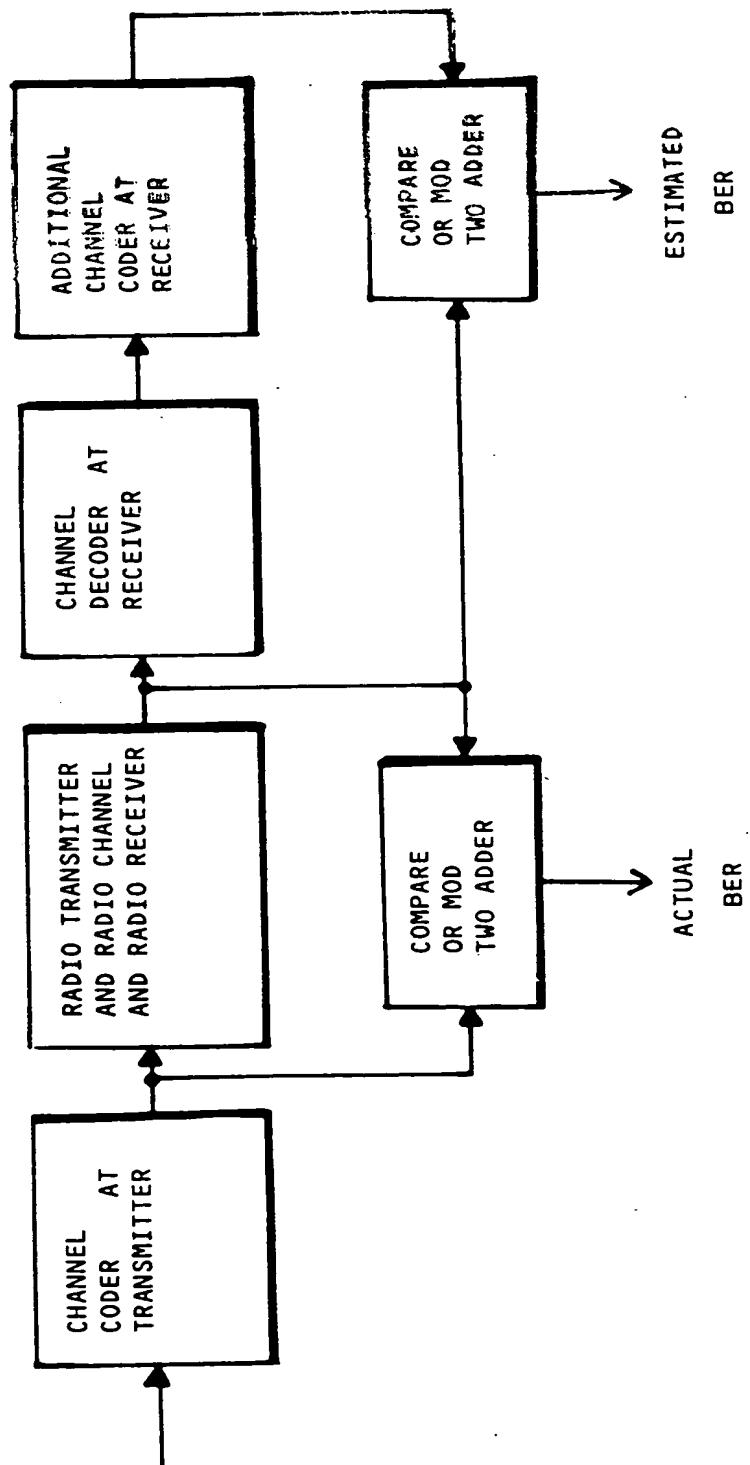


fig.7

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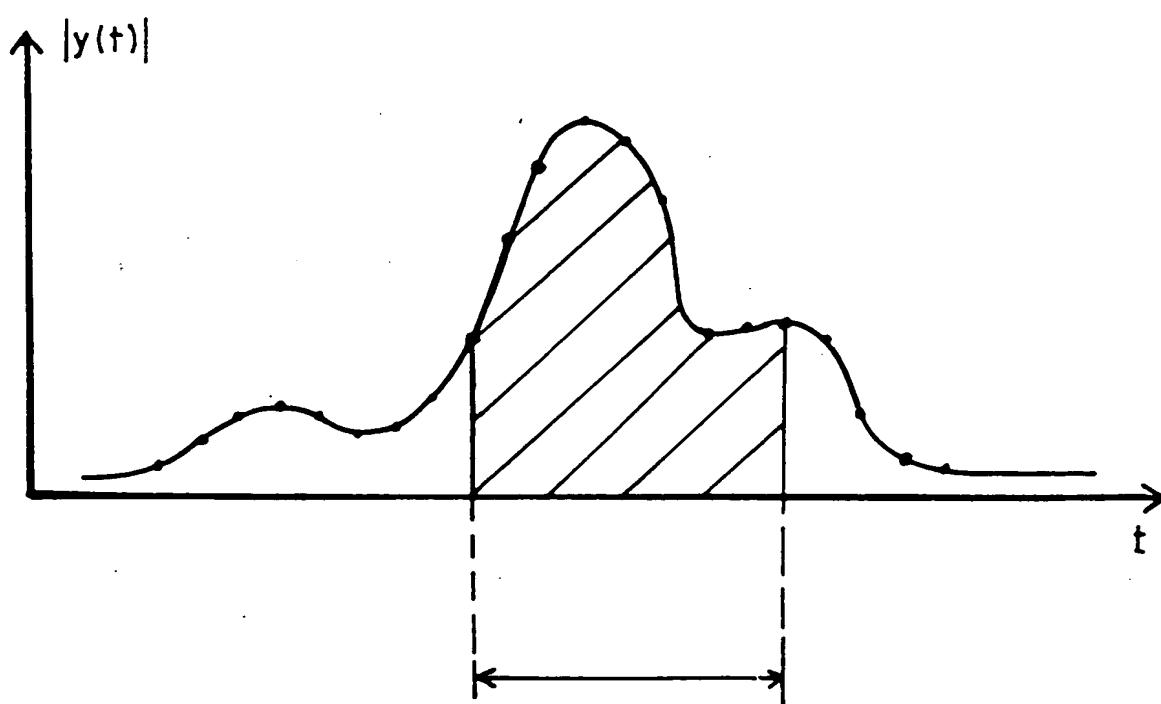


Fig.8

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INTERNATIONAL SEARCH REPORT

International Application No PCT/SE 90/00199

I. CLASSIFICATION OF SUBJECT MATTER (if several classification symbols apply, indicate all) ⁶ According to International Patent Classification (IPC) or to both National Classification and IPC IPC5: H 04 B 7/26, H 04 Q 7/04		
II. FIELDS SEARCHED		
Minimum Documentation Searched ⁷		
Classification System	Classification Symbols	
IPC5	H 04 B; H 04 Q	
Documentation Searched other than Minimum Documentation to the Extent that such Documents are Included in Fields Searched ⁸		
SE,DK,FI,NO classes as above		
III. DOCUMENTS CONSIDERED TO BE RELEVANT⁹		
Category	Citation of Document, ¹¹ with indication, where appropriate, of the relevant passages ¹²	Relevant to Claim No. ¹³
A	US, A, 4765753 (W.SCHMIDT) 23 August 1988, see column 6, line 66 - column 7, line 30 --	1-3
A	EP, A2, 0318033 (NEC CORPORATION) 31 May 1989, see column 7, line 52 - column 9, line 23 --	1
A	EP, A1, 0219559 (ANT NACHRICHTENTECHNIK GMBH) 29 April 1987, see column 13, line 11 - line 36 -- -----	1
<p>* Special categories of cited documents:¹⁰</p> <p>"A" document defining the general state of the art which is not considered to be of particular relevance</p> <p>"E" earlier document but published on or after the international filing date</p> <p>"L" document which may throw doubts on priority claim(s) or which is cited to establish the publication date of another citation or other special reason (as specified)</p> <p>"O" document referring to an oral disclosure, use, exhibition or other means</p> <p>"P" document published prior to the international filing date but later than the priority date claimed</p> <p>"T" later document published after the international filing date or priority date and not in conflict with the application but cited to understand the principle or theory underlying the invention</p> <p>"X" document of particular relevance, the claimed invention cannot be considered novel or cannot be considered to involve an inventive step</p> <p>"Y" document of particular relevance, the claimed invention cannot be considered to involve an inventive step when the document is combined with one or more other such documents, such combination being obvious to a person skilled in the art.</p> <p>"&" document member of the same patent family</p>		
IV. CERTIFICATION		
Date of the Actual Completion of the International Search	Date of Mailing of this International Search Report	
2nd July 1990	1990-07-10	
International Searching Authority	Signature of Authorized Officer	
SWEDISH PATENT OFFICE	GÖRAN MAGNUSSON <i>Goran Magnusson</i>	

ANNEX TO THE INTERNATIONAL SEARCH REPORT
ON INTERNATIONAL PATENT APPLICATION NO.PCT/SE 90/00199

This annex lists the patent family members relating to the patent documents cited in the above-mentioned international search report.
The members are as contained in the Swedish Patent Office EDP file on 90-05-24
The Swedish Patent Office is in no way liable for these particulars which are merely given for the purpose of information.

Patent document cited in search report	Publication date	Patent family member(s)	Publication date
US-A- 4765753	88-08-23	AU-B- 593619 AU-D- 6978287 DE-A- 3607687 EP-A- 0241954 JP-A- 62213435	90-02-15 87-09-10 87-09-10 87-10-21 87-09-19
EP-A2- 0318033	89-05-31	JP-A- 1180134 JP-A- 1183221 JP-A- 1141422	89-07-18 89-07-21 89-06-02
EP-A1- 0219559	87-04-29	NONE	